

503708370001

(19)



Europäisches Patentamt
European Patent Office
Office européen des brevets



(11)

EP 0 758 125 B1

(12)

EUROPEAN PATENT SPECIFICATION

(45) Date of publication and mention
of the grant of the patent:
23.01.2002 Bulletin 2002/04

(51) Int Cl.7: **G11B 20/10, G11B 19/12,**
G11B 19/26, G11B 23/36
// G11B7/00

(21) Application number: **96305781.5**

(22) Date of filing: **06.08.1996**

(54) **Recording medium, recording apparatus and reproducing apparatus**

Aufzeichnungsmedium, Aufzeichnungsgerät und Wiedergabegerät

Milieu d'enregistrement, appareil d'enregistrement et appareil de lecture

(84) Designated Contracting States:
AT DE FR GB NL

• **Ogura, Yasuhiro, c/o Sony Corp.**
Tokyo (JP)

(30) Priority: **09.08.1995 JP 22275095**
09.08.1995 JP 22274895

(74) Representative: **Nicholls, Michael John**
J.A. KEMP & CO. 14, South Square Gray's Inn
London WC1R 5JJ (GB)

(43) Date of publication of application:
12.02.1997 Bulletin 1997/07

(56) References cited:

(73) Proprietor: **SONY CORPORATION**
Tokyo (JP)

EP-A- 0 386 914	EP-A- 0 440 224
EP-A- 0 485 234	EP-A- 0 499 295
EP-A- 0 530 468	EP-A- 0 777 227
WO-A-93/07679	WO-A-96/30906
US-A- 5 315 571	

(72) Inventors:
• **Nishio, Ayataka, c/o Sony Corp.**
Tokyo (JP)

EP 0 758 125 B1

Note: Within nine months from the publication of the mention of the grant of the European patent, any person may give notice to the European Patent Office of opposition to the European patent granted. Notice of opposition shall be filed in a written reasoned statement. It shall not be deemed to have been filed until the opposition fee has been paid. (Art. 99(1) European Patent Convention).

Description

[0001] The present invention relates generally to a recording apparatus and a reproducing apparatus for a recording medium using a novel data format.

[0002] Recently, compact disks (CDS) have become highly popularized as recording media with high sound qualities. A compact disk is a disk-shaped recording medium having a diameter of approximately 12 cm, on which digital audio data is recorded, with a sampling frequency f_s selected to be 44.1 kHz and a quantizing bit number selected to be 16 bits.

[0003] Even more recently, various research has been performed so as to realize systems with higher sound qualities than those provided by the above-described CDS, which for the sake of easy explanation will be referred to as "the first generation CD", in connection with the development of various media having large storage capacities as well as high transfer rates. To realize such higher sound qualities, the first approach that was conceived is to increase the sampling frequency.

[0004] That is, when a sampling frequency f_s is equal to 44.1 kHz as in the first generation CD high-frequency components of the audio signal data would be limited to approximately 20 kHz. On the other hand, by setting the sampling frequency higher than 44.1 kHz audio data components higher than 20 kHz can be recorded in order that more naturally sounding audio signals are recorded/reproduced.

[0005] On the other hand, while it is possible to establish a novel CD media system with a data format whose sampling frequency is set to be higher than that of the first generation CD, there are various practical problems in this proposed CD media system.

[0006] Principally, even when the above-described CD media system with the novel format is realized, this novel CD media system should be compatible with the first generation CD in a practical sense.

[0007] For instance, a reproducing apparatus corresponding to the novel system CD with a data format whose sampling frequency is made higher than that of the first generation CD should also be able to reproduce the first generation CD. In such a case where disk compatibility is added to a reproducing apparatus, two separate reproducing-system circuits would at least be required in the digital section of the system, that is, a decoder and a D/A converter corresponding to the first generation CD would be required, and another decoder and D/A converter corresponding to the novel system CD would also be required. Moreover, separate clock generators would have to be separately employed in the respective reproducing system circuits.

[0008] This approach would make the circuit arrangement of the reproducing apparatus more complex and larger in scale, as well as raise costs both of which would be unacceptable.

[0009] A playback device with compatibility as described above has inevitably such structure as shown

in Fig. 6. In Fig. 6, the disk that is mounted on the turntable may be a first generation CD 1a or a second generation CD 1b. When the first generation CD 1a is mounted, a pickup 41 reads table of contents (TOC) information on the inner periphery of the disk while the disk 1 is being rotated by a spindle motor 46. A disk discriminating unit 42 judges whether the disk 1 is a first generation CD 1a or a second generation CD 1b based on the TOC information.

[0010] If the disk 1 is a first generation CD 1a, the disk discriminating unit 42 operates switches 51 and 52 to connect the output of each switch to the respective input terminal T1. An oscillator 43 generates a master clock signal CK1 having a prescribed frequency for processing the first generation CD 1a, the connection of the output of switch 51 to the input terminal T1 results in the supply of the master clock signal CK1 to a motor controller 45. The motor controller 45 generates the desired standard clock signal by frequency dividing the master clock signal CK1, and using the standard clock signal, the motor controller 45 controls the spindle motor 46 so that the disk 1 is rotated at a constant linear velocity. In this example, the constant linear velocity is that prescribed for the first generation CD 1a.

[0011] The master clock signal CK1 generated by the oscillator 43 is fed to a first generation CD decoder 47 and a D/A converter 49 for the first generation CD.

[0012] The decoder 47 for the first generation CD exercises RF processing, Eight-Fourteen demodulation, and error correction processing to the pit information read from the disk 1a by the pickup 41 and outputs digital audio data having a sampling frequency f_s of 44.1 kHz and a quantization number of 16 bits.

[0013] The digital audio data is converted to an analog audio signal by the D/A converter 49, and the analog audio signal is output at a terminal 53 through the input terminal T1 and the output of the switch 52.

[0014] When the second generation CD 1b is mounted, the disk discriminating unit 42 judges whether the disk 1 is a second generation CD based on the TOC information, and the outputs of switches 51 and 52 are connected to the respective input terminals T2.

[0015] An oscillator 44 generates a master clock signal CK2 having a prescribed frequency for processing the second generation CD 1b, and the connection of the output of switch 51 to the input terminal T2 results in the supply of the master clock signal CK2 to the motor controller 45. The motor controller 45 generates a desired standard clock by frequency dividing the master clock signal CK2, and using the standard clock signal the motor controller 45 controls the spindle motor 46 so that the disk 1 is rotated at a constant linear velocity, which in this example is the constant linear velocity prescribed for the second generation CD 1b.

[0016] The master clock signal CK2 generated by the oscillator is fed to a decoder 48 for the second generation CD and to a D/A converter 50. The decoder 48 for the second generation CD exercises RF processing,

Eight-Fourteen demodulation, and error correction processing to the pit information read from the disk 1 by the pickup 41 and outputs digital audio data having a sampling frequency f_s of 96 kHz and quantization number of 24 bits.

[0017] The digital audio data is converted to an analog audio signal by a second D/A converter 50, and the analog audio signal is output at terminal 53 through the input terminal T2 and the output of the switch 52 as the playback output.

[0018] In order to satisfy the compatibility requirement between the first generation CD 1a and the second generation CD 1b, the playback device is inevitably structured as shown in Fig. 6, however, as seen from the above compatibility requirement such results in complex circuit structure, large size, and expensive cost of the playback device, thus, this method is considered to be disadvantageous. The requirement of two sets of D/A converters, including peripheral devices of the D/A converter such as an amplifier and filter not shown in Fig. 6, is the largest cause of the complex circuit structure of the playback device.

[0019] There is another problem on the CD manufacturing side. For example, considering that both a novel type CD and a first generation CD were to be manufactured from a certain music source, such as a master tape of a musical composition and the like, the audio signal derived from the master tape is digitized to produce digital audio data for the novel type CD, while using, for instance, a sampling frequency of 96 kHz.

[0020] As a result, the novel type CD with the high sound quality is realized, but the first generation CD manufactured at the same time with the novel type CD would have the difficulty that the digital data with a sampling frequency of 96 kHz must be converted into digital data with a sampling frequency of 44.1 kHz. At this time, the sound quality of the first generation CD would deteriorate due to the jitter phenomenon caused by the sampling rate converting process.

[0021] WO-A-93/07679 discloses a receiver using a Sigma-Delta modulator ($\Delta\Sigma$ modulator). The input signal is A/D converted by the modulator to a 1-bit data stream and thereafter down-sampled to a 16-bit data stream.

[0022] EP 0,499,295 A discloses a CD in which digital audio data is stored at the inner periphery and an FM-modulated video signal is stored at an outer periphery.

[0023] The present invention has been made to alleviate these problems and, therefore, has as an object to provide a novel type recording medium capable of having better compatibility with the first generation recording medium, and also a recording apparatus and a reproducing apparatus that can employ such a novel type recording medium.

[0024] More specifically, as a recording medium wherein while a frequency n , where n is an integer, times 44.1 kHz is used as a sampling frequency, a digitized audio signal is recorded thereon. This digitized audio

signal is equal to 1 bit $\Delta\Sigma$ -modulated signal.

[0025] Accordingly, the present invention provides a recording apparatus for recording a first digital signal consisting of multi-bit data words sampled at a first sampling frequency to a first recording medium and a second digital signal consisting of 1 bit data words (data having a quantisation level of one bit) sampled at a second sampling frequency higher than the first sampling frequency to a second recording medium comprising:

analog to digital converting means for converting an input analog signal to a one-bit data stream signal (a data stream having a quantisation level of one bit) at the second sampling frequency;
down sampling and re-quantisation means for converting the one-bit data stream signal sampled at the second sampling frequency to a multi-bit digital signal sampled at the first sampling frequency;
first recording signal processing means for processing the multi-bit digital signal for recording;
first recording means for recording the digital signal processed by the first signal processing means to the first recording medium;
second recording signal processing means for processing the one-bit data stream signal for recording; and
second recording means for recording the digital signal processed by the second signal processing means to the second recording medium.

[0026] Further, the present invention provides reproducing apparatus as defined in claims 8 and 16.

[0027] The present invention can be used for a recording medium for recording digital audio data whose sampling frequency is n , where n is an integer, times 44.1 kHz, and also a recording apparatus and a reproducing apparatus corresponding to this recording medium. Since the sampling frequency is made higher than 44.1 kHz, a higher sound quality can be realized. In particular, since the recording data is $\Delta\Sigma$ -modulated with 1-bit quantization, the sampling frequency can be set to an even higher frequency, so that higher sound qualities can be sufficiently obtained.

[0028] Moreover, since the sampling frequency is an integer multiple of the 44.1 kHz employed in the presently available CD system, better matching characteristics between the second generation CD system and the presently available CD system are obtained. Especially, considering the compatibility of the recording medium according to the present invention with the presently available recording medium, since the sampling frequency for the recording medium is a multiple integer of that for the presently available recording medium, compatibility is realized with the present invention without an overly complex circuit arrangement for the reproducing apparatus. Also, as a sampling rate converter is not required, there is no deterioration in sound qualities caused by jitter components.

[0029] Similarly, in the recording apparatus, since it too has good matching characteristics, and since its sound quality is not deteriorated by jittering, both this recording medium and the presently available recording medium can be readily manufactured for the same sound source.

[0030] As a recording apparatus, there are provided digital audio signal producing means for digitizing an analog audio signal by using a sampling frequency n , where n is an integer, times 44.1 kHz to produce a digital audio signal, and recording means capable of performing preselected recording signal processing on the digital audio signal produced by the digital audio signal producing means to thereby record the processed digital audio signal on a recording medium. Thus, the above-described new recording medium is capable of being produced.

[0031] The recording apparatus is further comprised of a filter means for performing a decimation process for reducing the sampling frequency by $1/n$ with respect to the digital audio signal produced by the digital audio signal producing means, whereby a digital audio signal whose sampling frequency is 44.1 kHz is produced, and a second recording means capable of performing predetermined recording signal processing of the digital audio signal obtained by the filter means to thereby record the processed digital audio signal onto a recording medium. As a consequence, the first generation recording medium whose sampling frequency is 44.1 kHz may be manufactured without deterioration of the sound quality.

[0032] As a reproducing apparatus, there are provided clock generating means for generating a clock having a predetermined frequency, decoding means for extracting a digital audio signal from information read out from a recording medium based upon a sampling frequency n times 44.1 kHz and employing the clock output from the clock generating means, and D/A converting means for converting the digital audio signal obtained from the decoding means into an analog signal. Thus, it is possible to realize a reproducing apparatus corresponding to the above-described recording medium.

[0033] The reproducing apparatus is further comprised of dividing means for dividing the clock signal generated from the clock generating means into a second clock signal, and second decoding means for extracting a digital audio signal from the information read out from the recording medium based on a sampling frequency of 44.1 kHz with employment of the second clock signal. Accordingly, the recording apparatus is made compatible with the first generation recording medium without a massive increase in the scale of the circuit arrangement.

[0034] In accordance with another aspect a sampling frequency conversion means for converting digital data having a first sampling frequency output from a first decoding means for the first recording medium to the digital data having the second sampling frequency is provided. A D/A conversion means for operating on the out-

put generated from a second decoding means for the second recording medium and on the output generated from the sampling frequency conversion means is provided.

[0035] More specifically, the D/A conversion means is provided as a conversion means for converting digital data having the second sampling frequency to the analog audio signal, and in respect of the decoded digital data having the first sampling frequency, the data having the first sampling frequency is converted to the data having the second sampling frequency, then, the data having the second sampling frequency is fed to D/A conversion means to convert the digital data to the analog signal.

[0036] The invention will be further described by way of example with reference to the accompanying drawings, in which:-

[0037] Fig. 1 is a block diagram showing a recording apparatus according to an embodiment of the present invention.

[0038] Figs. 2A-2D are explanatory diagrams for explaining a frequency spectrum of a signal at each operating stage of recording apparatus according to the embodiment.

[0039] Fig. 3 is a block diagram showing a reproducing apparatus according to an embodiment of the present invention.

[0040] Fig. 4 is a block diagram showing a reproducing apparatus according to another embodiment of the present invention.

[0041] Fig. 5 is a block diagram showing a reproducing apparatus according to yet another embodiment of the present invention.

[0042] Fig. 6 is a block diagram showing a reproducing apparatus that was previously proposed.

[0043] Referring now to Figs. 1 to 5, embodiments according to the present invention will be explained.

[0044] As a recording medium in accordance with one embodiment of the present invention, a recording medium with a higher sound quality may be realized with respect to the first generation CD whose sampling frequency f_s is equal to 44.1 kHz and whose quantizing bit number is selected to be 16 bits. A sampling frequency of this novel recording medium is selected to be 2.8224 MHz, typically f_s is 44.1 kHz and 2.8224 MHz is indicated as 64 f_s hereinafter, which is 64 times the sampling frequency of the first generation CD. Furthermore, a digital audio signal of this 64 f_s sampling frequency, which has been modulated by 1-bit $\Delta\Sigma$ modulation, is employed as a digital audio signal to be recorded. Such a recording medium (CD) will be referred to hereinafter as a second generation CD. The above-described second generation CD has a diameter of approximately 12 cm, the same as that of the first generation CD.

[0045] The second generation CD has a diameter of 12 cm, a disk thickness of 1.2 mm, and a track pitch of 0.74 μm . The laser wavelength is selected to be either 650 nm or 635 nm, which is shorter than that for the

conventional or first generation, CD. Thus, the recording density of the second generation CD is increased to 4.7 Gigabytes to achieve a higher density disk.

[0046] Furthermore, another second generation CD has been proposed, in which the disk thickness of this second generation CD is subdivided into two thicknesses, that is, 0.6 mm and 0.6 mm, so that two recording layers are provided on one surface of the disk. The track pitch is selected to be 0.74 μm , so that the recording capacity of each recording layer is approximately 4.7 Gigabytes and, therefore, the entire recording capacity of this second generation CD becomes approximately 8.5 Gigabytes in total.

[0047] In addition to the above-explained disk having the two recording layers on one surface thereof, a still further second generation CD has been proposed, in which recording regions are formed on both surfaces of this disk which has a diameter of 12 cm and a disk thickness of 1.2 mm, and the track pitch thereof is 0.74 μm . Accordingly, this double recording surface disk has a recording capacity of 9.4 Gigabytes in total.

[0048] As to the 1-bit $\Delta\Sigma$ -modulated signal, a sampling frequency thereof may be set to achieve considerably higher values of data capacity and data transfer rate as compared with conventional PCM-modulated signals. Therefore, in accordance with this embodiment of the present invention, while the sampling frequency is selected to be 64 fs, it is possible that audio data containing a high frequency component up to 1.4 MHz may be recorded/reproduced. As a result, it is possible to provide a second generation CD whose sound qualities are extremely improved compared with those of the first generation CD.

[0049] Also, since the sampling frequency of the second generation CD is an integer multiple of the sampling frequency of the first generation CD, this second generation CD matches the first generation CD system and compatibility is maintained.

[0050] First, a description will be made of a recording apparatus corresponding to this second generation CD with reference to Figs. 1 and 2A-2D.

[0051] Fig. 1 is a schematic block diagram of a recording apparatus that can be used in the manufacture of the second generation CD, as well as the first generation CD.

[0052] An analog audio signal is fed in at input terminal 10 as an original audio signal derived from a master tape. A frequency spectrum of the analog audio signal is represented in Fig. 2A.

[0053] This analog audio signal is converted into digital data by a 1-bit $\Delta\Sigma$ modulation A/D converter 11. At this time, because the sampling frequency is selected to be 64 fs a digital audio signal with a 64 fs/1-bit format is produced. A frequency spectrum of this 64 fs/1-bit formatted digital audio signal is shown in Fig. 2B. In other words, in principle, audio data in a frequency range up to 32 fs may be digitized and substantially all signal components of the analog audio signal shown in Fig. 2A re-

main as digital data.

[0054] Also, quantizing noise components appear under a condition such that they are collected on a high frequency side of a frequency range due to the noise shaping function achieved in the $\Delta\Sigma$ modulation.

[0055] This 64 fs/1-bit digital audio signal is supplied to a recording signal processing unit 17 so as to be directly modulated to produce a recording signal. That is, the 64 fs/1-bit signal is directly used as the digital audio data to be recorded. In the recording signal processing unit 17, for instance, an error correction code is added to the data and also a modulating process such as an Eight-Fourteen Modulation (EFM) for recording purposes is performed. Then, a signal corresponding to the actual pit information to be formed on a disk is produced as a recording signal by this recording signal processing unit 17.

[0056] This recording signal is supplied to a disk recorder 18 so as to perform a recording operation based on the recording signal, namely, a pit forming operation for an original disk, so that an original CD comprising music software may be manufactured as the second generation CD 1b. From this original CD 1b manufactured in the above-described manner, a stamper (not shown) is fabricated, whereby large numbers of the second generation CD 1b may be mass produced using the stamper.

[0057] In the case where a first generation CD is also manufactured for the same music software, the 64 fs/1-bit signal from the $\Delta\Sigma$ -modulation 1-bit A/D converter 11 is supplied to a first decimation filter 12 so as to be converted into 2fs (=88.2 kHz)/24-bit digital data. A frequency spectrum of the 2fs/24-bit digital data is shown in Fig. 2C. That is, since the sampling frequency is reduced to 2fs, a data component in a frequency range up to the frequency fs is left.

[0058] This 2fs/24-bit digital data is filtered by a second decimation filter 13 to become fs(=44.1 kHz)/24-bit digital data. A frequency spectrum of the fs/24-bit digital data is shown in Fig. 2D. A data component of a frequency range up to (1/2) fs remains.

[0059] The sampling frequency is set to 1/64 by way of these decimation filters 12 and 13. This process operation does not correspond to a so-called sampling rate conversion but, rather, is due to perfectly synchronized digital filters for performing the 64:1 decimation. Accordingly, there is no factor present to cause a jitter component.

[0060] This fs/24-bit digital data from the decimation filter 13 is converted by a bit mapping unit 14 into data whose quantizing bit number is 16 bits, which will then be supplied to a recording signal processing unit 15. In the recording signal processing unit 15, a preselected process operation such as adding an error correction code and performing Eight-Fourteen Modulation (EFM) is carried out on the fs/16-bit digital audio signal so as to thereby produce a recording signal. This recording signal is supplied to another disk recorder 16. Based on

this recording signal, the disk recorder 16 forms pits in an original disk, so that an original CD of the music software is formed as a first generation CD 1a. A stamper (not shown) is fabricated from this original CD 1a manufactured in this manner. Accordingly, the first generation CD 1a is then mass produced using the stamper.

[0061] As described above, in accordance with the recording apparatus of Fig. 1, the first generation CD 1a and the second generation CD 1b can be manufactured. Here, since the 64 fs/1-bit digital audio data is directly employed as the recording signal for the second generation CD, this second generation CD 1b becomes a disk whose sound quality is considerably improved.

[0062] Also, since the sampling frequency of the second generation CD 1b is an integer multiple of that of the first generation CD, the sound quality of the fabricated first generation CD 1a is not deteriorated compared with that of a conventional first generation CD. In other words, in performing the sampling frequency conversion only a filtering process for performing the 1/64 decimation is required and, because the sampling rate converter is no longer required, the jitter components that accompany sampling rate conversion are not produced. As a consequence, it is possible to achieve a sound quality of this manufactured first generation CD 1a that is equivalent to that of a conventional CD obtained when the analog audio signal is directly sampled by 44.1 kHz.

[0063] Next, a description is made of a reproducing apparatus corresponding to the second generation CD 1b.

[0064] Fig. 3 is a schematic block diagram showing major portions of the reproducing apparatus. It should be noted that this reproducing apparatus is also capable of reproducing the first generation CD 1a. A disk 1, which can be either the first generation CD 1a or the second generation CD 1b, is rotatably driven by a spindle motor 26.

[0065] The spindle motor 26 is driven in a constant linear velocity (CLV) mode in response to a drive signal from a motor controller 25.

[0066] A detailed description of spindle servo operation by the motor controller 25 for CLV control is omitted, because such servo systems are well known. A clock signal CK2 produced by an oscillator 23 is divided by a divider 24 to thereby obtain a predetermined reference clock signal CKs. Then, a PLL system clock that has been synchronized with the reproduced data is compared in the motor controller 25 with this reference clock CKs so as to produce an error signal. Power is supplied to the spindle motor 26 according to this error signal, so that the CLV servo is operated. Although not shown in the drawing, the PLL system clock may be implemented, for instance, by supplying extracted data into the PLL circuit within a first generation CD decoder 28 and a second generation CD decoder 29.

[0067] While the disk 1 is rotated, a pickup 21 irradiates laser light onto the recording surface of the disk 1, and then the laser light reflected from the recording sur-

face thereof is detected by the pickup 21, so that the information formed on the disk 1 in the form of pits may be read out.

[0068] The information read by the pickup 21 is supplied to a first generation CD decoder 28 and to a second generation CD decoder 29.

[0069] A clock signal CK2 having a frequency that will be used to decode the data in the second generation CD decoder 29 is generated by an oscillator 23, and this clock signal CK2 is supplied to the second generation CD decoder 29. The clock signal CK2 generated by the oscillator 23 is also divided by another divider 27 to produce another clock signal CK1 having a frequency that will be used to decode the data in the first generation CD decoder 28, and this second clock signal CK1 is supplied to the first generation CD decoder 28.

[0070] A 16-bit digital audio signal whose sampling frequency is equal to fs is decoded and output from the first generation CD decoder 28. This 16-bit signal is then converted into a 1-bit digital audio signal whose sampling frequency is equal to 64 fs by way of an oversampling filter 30 and a $\Delta\Sigma$ modulation circuit 31. Then, this 1-bit digital audio signal is supplied to an input terminal T1 of a switch 32.

[0071] The output from the switch 32 is supplied to a 1-bit D/A converter 33 to be converted into an analog audio signal, which is then output from an output terminal 34. To this 1-bit D/A converter 33, the clock signal CK2 generated by the oscillator 23, namely, a clock signal identical to the clock signal for the second generation CD decoder 29, is supplied.

[0072] A disk discriminating unit 22 functions to discriminate whether the loaded disk 1 is a first generation CD 1a or a second generation CD 1b. This discrimination may be achieved by reading the table of contents (TOC) data recorded on the innermost peripheral side of the disk.

[0073] The disk discriminating unit 22 controls operation of the switch 32, as well as the dividing ratio of the divider 24 according to the discrimination result.

[0074] Operations of the above-described reproducing apparatus will now be explained.

[0075] First, it is assumed that the disk 1 to be reproduced is a second generation CD 1b. When the disk discriminating unit 22 initially discriminates the loaded disk 1 as a second generation CD 1b based upon the TOC data, the disk discriminating unit 22 sets the dividing ratio of the divider 24 to the proper value for a second generation CD 1b. Also, the disk discriminating unit 22 causes the output of the switch 32 to be connected to a terminal T2.

[0076] Since the dividing ratio of the divider 24 is set as the value corresponding to the second generation CD 1b, the frequency of the reference clock signal CKs used for the CLV servo in the motor controller 25 becomes the frequency corresponding to the second generation CD 1b. That is, the disk 1 is rotatably driven at a linear speed corresponding to the second generation CD 1b.

[0077] At this time, since the pit information extracted by the pickup 21 is decoded by the second generation CD decoder 29, the 64 fs/1-bit digital audio signal is decoded and since the output of the switch 32 is connected to the terminal T2, the 64 fs/1 bit digital audio signal is supplied to the 1-bit D/A converter 33 so as to be thereby converted into the analog audio signal.

[0078] Secondly, it is assumed that the disk 1 to be reproduced is a first generation CD 1a, so that when the disk discriminating unit 22 initially discriminates the loaded disk 1 as a first generation CD 1a based upon the TOC data, the disk discriminating unit 22 sets the dividing ratio of the divider 24 to a proper value for a first generation CD 1a. Also, the disk discriminating unit 22 causes the output of the switch 32 to be connected to the input terminal T1.

[0079] Since the dividing ratio of the divider 24 is set to the value corresponding to a first generation CD 1a, the frequency of the reference clock signal CKs used for the CLV servo in the motor controller 25 becomes the frequency corresponding to a first generation CD 1a. That is, the disk 1 is rotatably driven at a linear speed corresponding to a first generation CD 1a.

[0080] At this time, since the pit information extracted by the pickup 21 is decoded by the first generation CD decoder 28, the fs/16-bit digital audio signal is properly decoded. This fs/16-bit digital audio signal is processed by the over sampling filter 30 and the $\Delta\Sigma$ modulation circuit 31, which are operated in response to the clock signal CK2, to be converted into a 64 fs/1 bit digital audio signal. Then, since the output of the switch 32 is connected to the terminal T1, the 64 fs/1 bit digital audio signal is supplied to the 1 bit D/A converter 33 to be thereby converted into the analog audio signal.

[0081] By reproducing the second generation CD 1b in accordance with such a reproducing apparatus, audio data with very high sound quality can be reproduced using 64fs. Also, since the sampling frequency of the second generation CD 1b is an integer multiple of the sampling frequency of the first generation CD 1a, as represented in Fig. 3, it is possible to realize a reproducing apparatus having compatibility, even when the clock system and the reproduction system do not involve very complex circuit arrangements.

[0082] In other words, as to the clock system, since the ratio of the sampling frequency of the first generation CD 1a to that of the second generation CD 1b constitutes an integer ratio, the clock signal from the oscillator 23 can be used in common. That is to say, even when a plurality of oscillators are not employed, it is possible to easily generate clocks having the necessary frequencies by using frequency dividers. As a result, two independent master clock systems need not be constructed, and the circuit arrangement of the clock system can be made simple.

[0083] Also, with respect to the reproducing system, the 1-bit D/A converter 33 can be used in common, so that the circuit arrangement of the reproducing system

can also be made simple. Moreover, sound quality does not deteriorate using this system.

[0084] Since the 1-bit D/A converter 33 is a D/A converter that performs operations according to data reproduced from the second generation CD 1b, there are absolutely no problems with respect to a second generation CD 1b. To also use this 1-bit D/A converter 33 for the data reproduced from the first generation CD 1a, fs/16-bit data decoded from the first generation CD decoder 28 must be converted into 64 fs/1-bit data. Since the sampling frequency is an integer multiple, however, the fs/16-bit data is simply oversampled 64 times by the oversampling filter 30 and converted into 64 fs/1-bit data by the $\Delta\Sigma$ modulation circuit 31. Thus, a sampling rate converter is not required. As a consequence, since there is no factor to produce the jitter component, the sound quality of the first generation CD 1a is not deteriorated.

[0085] Fig. 4 represents another arrangement of the reproducing apparatus, in which the same reference numerals shown in Fig. 3 are employed to denote the same functional units and explanations thereof are omitted.

[0086] In this case, a D/A converter 36 corresponds to a D/A converter operated in response to the clock signal CK1 and has a sampling frequency fs.

[0087] As a result, when the first generation CD 1a is reproduced the fs/16-bit reproduced data output from the first generation CD decoder 28 is directly supplied via the switch 32 to the D/A converter 36, so that this reproduced data is converted into analog audio data.

[0088] on the other hand, when the second generation CD 1b is reproduced, the 64fs/1-bit reproduced data outputted from the second generation CD decoder 29 is processed in a decimation filter 35 using 64:1 decimation processing. Then, since this fs/16-bit reproduced data is supplied via the switch 32 to the A/D converter 36, this fs/16-bit reproduced data is converted into analog audio data.

[0089] In this case, as with the reproducing apparatus of Fig. 3, compatibility can be realized by a simple clock reproducing system, and there is no deterioration in the sound quality caused by joint use of the D/A converter 36. That is, there is no possibility that jitter components are produced in the decimation process of the decimation filter 35.

[0090] Fig. 5 represents another arrangement of the reproducing apparatus, in which the same reference numerals shown in Figs. 3 and 4 are employed to denote the same functional units and explanations thereof are omitted.

[0091] Although the description of the spindle servo operation of the constant linear velocity control by the motor controller 25 is not repeated, the clock signal CK1 generated by a first oscillator 60 or the clock signal CK2 generated by a second oscillator 61 is fed to the spindle motor 26 through the output of a switch 62. The clock signal CK1 or clock signal CK2 is subjected to a required frequency division processing, and the standard clock signal having a prescribed frequency corresponding to

the first generation CD 1a or the second generation CD 1b is compared with PLL clock synchronous with the playback data to generate an error signal. Then, the control linear velocity servo is exercised by applying power to the spindle motor spindle motor depending on the error signal. Although the PL clock is not shown in the drawings, the PL clock signal is generated by feeding the data extracted in the decoder 28 for the first generation CD when the first generation CD 1a is regenerated or from the decoder 29 for the second generation CD when the second generation CD 1b is regenerated to the PLL circuit.

[0092] When the disk 1 is rotated, the pickup 21 irradiates a laser beam on the record surface of the disk 1, and the reflected beam is detected to read the pit information formed on the disk 1.

[0093] The information read by the pickup 21 is fed to the decoder 28 for the first generation CD and to the decoder 29 for the second generation CD. The decoder 28 for the first generation CD and the decoder 29 for the second generation CD each exercise RF processing, Eight-Fourteen demodulation, error correction, and de-interleave processing depending on the first generation CD 1a and the second generation CD 1b, respectively, to obtain digital audio data.

[0094] The clock signal is fed from the oscillator 60 to the decoder 28 for the first generation CD as a clock signal for the decoding processing. The clock signal CK2 is fed from a second oscillator 61 to the decoder 29 for the second generation CD as a clock signal for the decoding processing. The digital audio signal having a sampling frequency f_s of 44.1 kHz and a quantization bit number of 16 bit is output from the decoder 28 for the first generation CD as the decoded output. The digital audio signal having a sampling frequency f_s of 96 kHz and a quantization bit number of 24 bits is output from the decoder 29 for the second generation CD as the decoded output.

[0095] The output from the decoder 28 for the first generation CD is fed to a sampling rate converter 63. The clock signal CK1 and the clock signal CK2 are fed to the sampling rate converter 63. The sampling rate converter 63 converts the digital audio signal having a sampling frequency f_s of 44.1 kHz and 16 quantization bits to a digital audio signal having a sampling frequency f_s of 96 kHz and 24 quantization bits.

[0096] The output from the sampling rate converter 63 and the output from the decoder 29 for the second generation CD are fed to a D/A converter 64 selectively by switch 65. The clock signal CK2 is fed to the D/A converter 64 as a clock signal for the processing, the D/A converter 64 converts the digital audio signal having a sampling frequency f_s of 96 kHz and 24 quantization bits to the analog audio signal. The analog audio signal is outputted from the terminal 66 as the playback signal.

[0097] The disk discriminating unit 22 is a section for judging whether the disk 1 mounted is a first generation CD 1a or a second generation CD 1b. The judgment is

possible by reading TOC data recorded on the innermost periphery of the disk. The disk judging section 22 controls the switches 62 and 65 depending on the judged result.

[0098] Operation of such a playback device is described herein under.

[0099] First, operations for playback of the disk 1 of the second generation CD 1b are described. The disk discriminating unit 22 judges whether the disk 1 is a second generation CD 1b based on the TOC data on the disk 1, then, the disk discriminating unit 22 operates the switches 62 and 65 to connect the outputs thereof to the respective input terminals T2.

[0100] Thereby, the clock signal CK2 is fed to the motor controller 25, and the frequency for the standard clock signal used for the constant linear velocity servo is converted to the frequency which corresponds to the second generation CD 1b. In other words, the disk 1 is driven rotationally at a linear velocity which corresponds to the second generation CD 1b.

[0101] Then, the pit information extracted by the pickup 21 is decoded by the decoder 29 for the second generation CD, thereby, the digital audio signal having a sampling frequency f_s of 96 kHz and 24 quantization bits is obtained. When, the switch 65 is connected to the terminal T2, therefore, the digital audio signal of 96 kHz/24-bit format is fed to the D/A converter 64 and converted to the analog audio signal.

[0102] Next, operations for playback when the disk 1 is a first generation CD 1a. The disk discriminating unit 22 judges whether the disk 1 is a first generation CD 1a based on the TOC data on the disk 1, then, the disk discriminating unit 22 operates the switches 62 and 65 to connect the outputs thereof to the respective input terminals T1.

[0103] Thereby, the clock signal CK1 is fed to the motor controller 25. The frequency of the standard clock signal used for the constant linear velocity servo, which is obtained by frequency dividing the clock signal CK1, is converted to the frequency which corresponds to the first generation CD 1a. In other words, the disk 1 is driven rotationally at a linear velocity which corresponds to the first generation CD 1a.

[0104] Then, the pit information extracted by the pickup 21 is decoded by the decoder 28 for the first generation CD, thereby, the digital audio signal of 44.1 kHz/16-bit format is decoded.

[0105] The digital audio signal of 44.1 kHz/16-bit format is converted to the digital audio signal of 96 kHz/24-bit format by the sampling rate converter 29. Then, the output of switch 65 is connected to the input terminal T1, therefore, the digital audio signal of 96 kHz/24-bit format is fed to the D/A converter 64 and converted to the analog audio signal.

[0106] By using such a playback device, when a second generation CD 1b is regenerated, the audio data of 96 kHz/24-bit format is regenerated to provide high quality sound.

[0107] For playback of the first generation CD 1a, the decoded digital audio signal of 44.1 kHz/16-bit format is converted to the digital audio signal of 96 kHz/24-bit format by the sampling rate converter 63.

[0108] Therefore, the D/A converter 64 which corresponds to the digital audio signal of 96 kHz/24-bit format is used also for playback of the first generation CD 1a, thereby only one set of D/A converting circuits (D/A converter, and peripheral devices such as an amplifier and filter) is sufficient for the playback circuit, thus, the structure of the playback circuit is simplified.

[0109] It should be noted that although the above-explained embodiments have been described such that the presently available CD system is the first generation CD 1a and the second generation CD 1b has characteristics that match those of this first generation CD 1a, the present invention is not limited to CD systems.

[0110] For instance, in a digital tape recorder system it is possible to realize such a recording/reproducing system by employment of a sampling frequency that is an integer multiple of 44.1 kHz.

[0111] Also, in a recording/reproducing system whose sampling frequencies are selected to be 32 kHz and 48 kHz, when the present invention is applied, such a system may be constructed whose sampling frequency is set to either 32 kHz x n or 48 kHz x n, where n is an integer. In particular, a sampling frequency of 48 kHz x n may be suitably used in next-generation video recording/reproducing media.

Claims

1. A recording apparatus for recording a first digital signal consisting of multi-bit data words sampled at a first sampling frequency (fs) to a first recording medium (1a) and a second digital signal consisting of 1 bit data words sampled at a second sampling frequency (64fs) higher than the first sampling frequency (fs) to a second recording medium (1b) comprising:

analog to digital converting means (11) for converting an input analog signal to a one-bit data stream signal at the second sampling frequency (64fs);
down sampling and re-quantisation means (12, 13, 14) for converting the one-bit data stream signal sampled at the second sampling frequency (64fs) to a multi-bit digital signal sampled at the first sampling frequency (fs);
first recording signal processing means (15) for processing the multi-bit digital signal for recording;
first recording means (16) for recording the digital signal processed by the first signal processing means (15) to the first recording medium (1a);

second recording signal processing means (17) for processing the one-bit data stream signal for recording; and

second recording means (18) for recording the digital signal processed by the second signal processing means (17) to the second recording medium (16).

2. The recording apparatus according to claim 1, wherein said analog to digital converting means (11) comprises a one-bit delta-sigma modulation analog to digital converter.
3. The recording apparatus according to claim 1 or 2, wherein said down sampling and requantisation means (12, 13, 14) comprises a decimation filter (13) and bit-mapping device (14) in series with each other.
4. The recording apparatus according to claim 1, 2 or 3, wherein said first recording signal processing means (15) includes means for recording the digital signal formatted by 16-bit data words sampled at 44.1 kHz on the first recording medium (1a).
5. The recording apparatus according to claim 1, 2, 3 or 4, wherein said second recording signal processing means (17) includes means for recording the digital signal formatted by 1-bit data words sampled at 2.8224 MHz on the second recording medium (1b).
6. The recording apparatus according to claim 5, adapted to record said digital signal on said second recording medium (1b) having a base plate with a diameter not larger than 120 mm and a thickness larger than 0.6 mm.
7. The recording apparatus according to claim 6, for recording tracks on the second recording medium (1b) at a track pitch of approximately 0.74 μm , at an information recording capacity of not less than 4.7 Gbytes.
8. A reproducing apparatus for reproducing a first digital signal formatted by multi-bit data words sampled at a first sampling frequency (fs) from a first recording medium (1a) and a second digital signal formatted by 1-bit data words sampled at a second sampling frequency (64fs) higher than the first sampling frequency (fs) from a second recording medium (1b), comprising:

a pick-up head (21) for reproducing a digital signal from one of the first and second recording media (1a, 1b);
first decoding means (28) for decoding the multi-bit data words sampled at the first sampling

frequency (f_s) corresponding to the first recording medium (1a);
 second decoding means (29) for decoding the 1-bit data words sampled at the second sampling frequency (64fs) corresponding to the second recording medium (1b);
 discriminating means (22) for discriminating the first recording medium (1a) and the second recording medium (1b); and
 switching means (65) for switching digital output signals from the first decoding means (28) and the second decoding means (29) in accordance with a result of said discriminating means (22).

9. The reproducing apparatus according to claim 8, further comprising:

over sampling means (63) for converting the sampling frequency of the digital output signal from the first decoding means (28) to the second sampling frequency (64fs);
 delta-sigma modulating means for converting multi-bit data words sampled at the first sampling frequency (f_s) to 1-bit data words sampled at the second sampling frequency (64fs).

10. The reproducing apparatus according to claim 8 or 9, further comprising:

down sampling means for converting the sampling frequency of the digital signal from the second decoding means (29) to the first sampling frequency (f_s).

11. The reproducing apparatus according to claims 8, 9 or 10, wherein said first decoding means (28) decodes signals from said first recording medium (1a) formatted by 16-bit data words sampled at 44.1 KHz.

12. The reproducing apparatus according to claim 8, 9, 10 or 11, wherein said second decoding means (29) decodes signals from said second recording medium (1b) formatted by 1-bit data words sampled at 2.8224 MHz.

13. The reproducing apparatus according to claim 12, adapted for reproducing said second digital signal from said second recording medium (1b) having a base plate with a diameter not larger than 120 mm and a thickness larger than 0.6 mm.

14. The reproducing apparatus according to claim 13, for reproducing tracks from the second recording medium (1b) formatted at a track pitch of approximately 0.74 μm and with an information recording capacity of not less than 4.7 Gbytes.

15. The reproducing apparatus according to any one of claims 8 to 14, for reproducing a recording medium (1b) recorded with a high-speed one-bit data stream output from a delta-sigma modulation analog to digital converter at a sampling frequency of $n \times 44.1$ KHz, where n is an interger.

16. A reproducing apparatus for reproducing a first digital signal consisting of first multi-bit data words sampled at a first sampling frequency (f_s) from a first recording medium (1a) and a second digital signal consisting of a second multi-bit data word sampled at a second sampling frequency (64fs) higher than the first sampling frequency (f_s) from a second recording medium (1b), comprising:

reproducing means (21) for reproducing the digital signal from one of the first recording medium (1a) and the second recording medium (1b);
 discrimination means (22) for discriminating the first recording medium (1a) and the second recording medium (1b);
 first decoding means (28) for decoding the digital data from the first recording medium (1a);
 second decoding means (29) for decoding the digital data from the second recording medium (1b);
 first clock generating means (60) for generating a first clock signal (CK1) based upon a rotating drive signal for the first recording medium (1a);
 second clock generating means (61) for generating a second clock signal (CK2) based upon a rotating drive signal for the second recording medium (1b);
 switching means (62) for switching the first and second clock signals (CK1, CK2) in accordance with result of discrimination means (22); and
 rotating means (26) for rotating the recording medium (1a, 1b) in accordance with the clock signal output from the switching means (62).

17. The reproducing apparatus according to claim 16, further comprising:

sampling rate converting means (63) for converting a bit number and a sampling frequency of the digital signal from the first decoding means (1a); and
 second switching means (65) for switching the output signal from the sampling rate converting means (63) and the digital signal from the second decoding means (29).

18. The reproducing apparatus according to claim 16, further comprising:

second switching means (65) for switching the output signal from the first decoding means (28)

and the output signal from the second decoding means (29) to form a playback signal output from the apparatus.

Patentansprüche

1. Aufzeichnungsgerät zum Aufzeichnen eines ersten Digitalsignals, welches aus Multibit-Datenwörtern besteht, die mit einer ersten Abtastfrequenz (f_s) abgetastet werden, auf einem ersten Aufzeichnungsträger (1a), und eines zweiten Digitalsignals, welches aus 1-Bit-Datenwörtern besteht, die mit einer zweiten Abtastfrequenz ($64f_s$) abgetastet werden, die höher ist als die erste Abtastfrequenz (f_s), auf einem zweiten Aufzeichnungsträger (1b), mit:

einer Analog-Digital-Umsetzungseinrichtung (11) zum Umsetzen eines gelieferten Analogsignals in ein Ein-Bit-Datenstromsignal mit der zweiten Abtastfrequenz ($64f_s$);
einer Abwärtsabtastungs- und Requantisierungseinrichtung (12, 13, 14) zum Umsetzen des Ein-Bit-Datenstromsignals, welches mit der zweiten Abtastfrequenz ($64f_s$) abgetastet wird, in ein Multibit-Digitalsignal, welches mit der ersten Abtastfrequenz (f_s) abgetastet wird;
einer ersten Aufzeichnungssignal-Verarbeitungseinrichtung (15) zum Verarbeiten des Multibit-Digitalsignals zur Aufzeichnung;
einer ersten Aufzeichnungseinrichtung (16) zum Aufzeichnen des durch die erste Signalverarbeitungseinrichtung (15) verarbeiteten Digitalsignals auf dem ersten Aufzeichnungsträger (1a);
einer zweiten Aufzeichnungssignal-Verarbeitungseinrichtung (17) zum Verarbeiten des Ein-Bit-Datenstromsignals zur Aufzeichnung; und
einer zweiten Aufzeichnungseinrichtung (18) zum Aufzeichnen des durch die zweite Signalverarbeitungseinrichtung (17) verarbeiteten Digitalsignals auf dem zweiten Aufzeichnungsträger (1b).

2. Aufzeichnungsgerät nach Anspruch 1, wobei die Analog-Digital-Umsetzungseinrichtung (11) einer Ein-Bit-Delta-Sigma-Modulation-Analog-Digital-Umsetzer umfaßt.
3. Aufzeichnungsgerät nach Anspruch 1 oder 2, wobei die Abwärtsabtastungs- und Requantisierungseinrichtung (12, 13, 14) ein Dezimierungsfiler (13) und eine Einrichtung (14) zum Erfassen von Bits in Reihe miteinander umfaßt.
4. Aufzeichnungsgerät nach Anspruch 1, 2 oder 3, wobei die erste Aufzeichnungssignal-Verarbeitungseinrichtung (15) eine Einrichtung zum Aufzeichnen

des Digitalsignals, welches durch 16-Bit-Datenwörter, die mit 44,1 kHz abgetastet sind, formatiert ist, auf dem ersten Aufzeichnungsträger (1a) aufweist.

5. Aufzeichnungsgerät nach Anspruch 1, 2, 3 oder 4, wobei die zweite Aufzeichnungssignal-Verarbeitungseinrichtung (17) eine Einrichtung zum Aufzeichnen des Digitalsignals, welches durch 1-Bit-Datenwörter formatiert ist, die mit 2,8224 MHz abgetastet sind, auf dem zweiten Aufzeichnungsträger (1b) aufweist.
6. Aufzeichnungsgerät nach Anspruch 5, welches ausgelegt ist, das Digitalsignal auf dem zweiten Aufzeichnungsträger (1b) aufzuzeichnen, der eine Basisplatte mit einem Durchmesser von nicht größer als 120 mm und eine Dicke größer als 0,6 mm aufweist.
7. Aufzeichnungsgerät nach Anspruch 6 zum Aufzeichnen von Spuren auf dem zweiten Aufzeichnungsträger (1b) mit einer Spurteilung von ungefähr 0,74 μm bei einer Informationsaufzeichnungskapazität von nicht weniger als 4,7 Gigabyte.
8. Wiedergabegerät zum Wiedergeben eines ersten Digitalsignals, welches durch Multidatenwörter formatiert ist, die mit einer ersten Abtastfrequenz (f_s) abgetastet werden, von einem ersten Aufzeichnungsträger (1a), und eines zweiten Digitalsignals, welches durch 1-Bit-Datenwörter formatiert ist, die mit einer zweiten Abtastfrequenz ($64f_s$) abgetastet werden, die höher ist als die erste Abtastfrequenz (f_s), von einem zweiten Aufzeichnungsträger (1b), mit:

einem Abtastkopf (21) zum Reproduzieren eines Digitalsignals vom ersten oder zweiten Aufzeichnungsträger (1a, 1b);
einer ersten Decodiereinrichtung (28) zum Decodieren der Multibit-Datenwörter, die mit der ersten Abtastfrequenz (f_s) abgetastet werden, die sich für den ersten Aufzeichnungsträger (1a) eignet;
einer zweiten Decodiereinrichtung (29) zum Decodieren der 1-Bit-Datenwörter, die mit der zweiten Abtastfrequenz ($64f_s$) abgetastet werden, die sich für den zweiten Aufzeichnungsträger (1b) eignet;
einer Unterscheidungseinrichtung (22) zum Unterscheiden des ersten Aufzeichnungsträgers (1a) und des zweiten Aufzeichnungsträgers (1b); und
einer Umschalteneinrichtung (65) zum Umschalten von digitalen Ausgangssignalen von der ersten Decodiereinrichtung (28) und der zweiten Decodiereinrichtung (29) gemäß dem Ergebnis der Unterscheidungseinrichtung (22).

9. Wiedergabegerät nach Anspruch 8, welches außerdem aufweist:

eine Überabtasteinrichtung (63) zum Umsetzen der Abtastfrequenz des digitalen Ausgangssignals von der ersten Decodiereinrichtung (28) auf die zweite Abtastfrequenz (64fs); eine Delta-Sigma-Modulationseinrichtung zum Umsetzen von Multibit-Datenwörtern, die mit der ersten Abtastfrequenz (fs) abgetastet werden, in 1-Bit-Datenwörter, die mit der zweiten Abtastfrequenz (64fs) abgetastet werden.

10. Wiedergabegerät nach Anspruch 8 oder 9, welches außerdem aufweist:

eine Abwärtsabtasteinrichtung zum Umsetzen der Abtastfrequenz des Digitalsignals von der zweiten Decodiereinrichtung (29) auf die erste Abtastfrequenz (fs).

11. Wiedergabegerät nach Anspruch 8, 9 oder 10, wobei die erste Decodiereinrichtung (28) Signale vom ersten Aufzeichnungsträger (1a) decodiert, die durch 16-Bit-Datenwörter, die mit 44,1 kHz abgetastet sind, formatiert sind.

12. Wiedergabegerät nach Anspruch 8, 9, 10 oder 11, wobei die zweite Decodiereinrichtung (29) Signale vom zweiten Aufzeichnungsträger (1b) decodiert, die durch 1-Bit-Datenwörter, die mit 2,8224 MHz abgetastet sind, formatiert sind.

13. Wiedergabegerät nach Anspruch 12, welches ausgebildet ist, um das zweite Digitalsignal vom zweiten Aufzeichnungsträger (1b) zu reproduzieren, der eine Basisplatte mit einem Durchmesser von nicht größer als 120 mm und eine Dicke hat, die größer ist als 0,6 mm.

14. Wiedergabegerät nach Anspruch 13 zum Wiedergeben von Spuren vom zweiten Aufzeichnungsträger (1b), die mit einer Spurtellung von ungefähr 0,74 µm formatiert sind, und mit einer Informationsaufzeichnungskapazität von nicht weniger als 4,7 Gigabyte.

15. Wiedergabegerät nach einem der Ansprüche 8 bis 14 zum Reproduzieren eines Aufzeichnungsträgers (1b), der mit einem Hochgeschwindigkeits-Ein-Bit-Datenstrom beschrieben ist, der von einem Delta-Sigma-Modulations-Analog-Digital-Umsetzer mit einer Abtastfrequenz $n \times 44,1$ kHz ausgegeben wird, wobei n eine ganze Zahl ist.

16. Wiedergabegerät zum Wiedergeben eines ersten Digitalsignals, welches aus ersten Multibit-Datenwörtern besteht, die mit einer ersten Abtastfrequenz (fs) abgetastet werden, von einem ersten

Aufzeichnungsträger (1a), und eines zweiten Digitalsignals, welches aus einem zweiten Multibit-Datenwort besteht, welches mit einer zweiten Abtastfrequenz (64fs) abgetastet wird, die höher ist als die erste Abtastfrequenz (fs), von einem zweiten Aufzeichnungsträger (1b), mit:

einer Wiedergabeeinrichtung (21) zum Wiedergeben des Digitalsignals vom ersten Aufzeichnungsträger (1a) oder vom zweiten Aufzeichnungsträger (1b);
einer Unterscheidungseinrichtung (22) zum Unterscheiden des ersten Aufzeichnungsträgers (1a) und des zweiten Aufzeichnungsträgers (1b);
einer ersten Decodiereinrichtung (28) zum Decodieren der Digitaldaten vom ersten Aufzeichnungsträger (1a);
einer zweiten Decodiereinrichtung (29) zum Decodieren der Digitaldaten vom zweiten Aufzeichnungsträger (1b);
einer ersten Takterzeugungseinrichtung (60) zum Erzeugen eines ersten Taktsignals (CK1) auf der Basis eines Drehansteuersignals für den ersten Aufzeichnungsträger (1a);
einer zweiten Takterzeugungseinrichtung (61) zum Erzeugen eines zweiten Taktsignals (CK2) auf der Basis eines Drehansteuersignals für den zweiten Aufzeichnungsträger (1b);
einer Umschalteneinrichtung (62) zum Umschalten des ersten und des zweiten Taktsignals (CK1, CK2) gemäß dem Ergebnis der Unterscheidungseinrichtung (22); und
einer Dreheinrichtung (26) zum Drehen des Aufzeichnungsträgers (1a, 1b) gemäß dem Taktsignal, welches von der Umschalteneinrichtung (62) ausgegeben wird.

17. Wiedergabegerät nach Anspruch 16, welches außerdem aufweist:

eine Absatzraten-Umsetzungseinrichtung zum Umsetzen einer Bitzahl und einer Abtastfrequenz des Digitalsignals von der ersten Decodiereinrichtung (1a), und
eine zweite Umschalteneinrichtung (65) zum Umschalten des Ausgangssignals von der Absatzraten-Umsetzungseinrichtung (63) und des Digitalsignals von der zweiten Decodiereinrichtung (29).

18. Wiedergabegerät nach Anspruch 16, welches außerdem aufweist:

eine zweite Umschalteneinrichtung (65) zum Umschalten des Ausgangssignals von der ersten Decodiereinrichtung (28) und des Ausgangssignals von der zweiten Decodiereinrichtung (29), um ein Wiedergabesignal, welches vom Gerät ausge-

ben wird, zu bilden.

Revendications

1. Appareil d'enregistrement pour enregistrer un premier signal numérique qui est constitué par des mots de données à multiples bits ou multibits qui sont échantillonnés à une première fréquence d'échantillonnage (f_s) sur un premier support d'enregistrement (1a) et un second signal numérique qui est constitué par des mots de données d'un bit qui sont échantillonnés à une seconde fréquence d'échantillonnage ($64f_s$) supérieure à la première fréquence d'échantillonnage (f_s) sur un second support d'enregistrement (1 b) comprenant :

un moyen de conversion analogique-numérique (11) pour convertir un signal analogique d'entrée selon un signal de train de données d'un bit à la seconde fréquence d'échantillonnage ($64f_s$) ;

un moyen de sous-échantillonnage et de requantification (12, 13, 14) pour convertir le signal de train de données d'un bit qui est échantillonné à la seconde fréquence d'échantillonnage ($64f_s$) selon un signal numérique multibit qui est échantillonné à la première fréquence d'échantillonnage (f_s) ;

un premier moyen de traitement de signal d'enregistrement (15) pour traiter le signal numérique multibit pour l'enregistrement ;

un premier moyen d'enregistrement (16) pour enregistrer le signal numérique qui est traité par le premier moyen de traitement de signal (15) sur le premier support d'enregistrement (1a) ;

un second moyen de traitement de signal d'enregistrement (17) pour traiter le signal de train de données d'un bit pour l'enregistrement ; et
un second moyen d'enregistrement (18) pour enregistrer le signal numérique qui est traité par le second moyen de traitement de signal (17) sur le second support d'enregistrement (1b).

2. Appareil d'enregistrement selon la revendication 1, dans lequel ledit moyen de conversion analogique-numérique (11) comprend un convertisseur analogique-numérique à modulation delta-sigma d'un bit.
3. Appareil d'enregistrement selon la revendication 1 ou 2, dans lequel ledit moyen de sous-échantillonnage et de requantification (12, 13, 14) comprend un filtre de décimation (13) et un dispositif de mise en carte de bits (14) en série l'un avec l'autre.
4. Appareil d'enregistrement selon la revendication 1,

2 ou 3, dans lequel ledit premier moyen de traitement de signal d'enregistrement (15) inclut un moyen pour enregistrer le signal numérique qui est formaté au moyen de mots de données de 16 bits échantillonnés à 44,1 kHz sur le premier support d'enregistrement (1a).

5. Appareil d'enregistrement selon la revendication 1, 2, 3 ou 4, dans lequel ledit second moyen de traitement de signal d'enregistrement (17) inclut un moyen pour enregistrer le signal numérique qui est formaté au moyen de mots de données d'un bit échantillonnés à 2,8224 MHz sur le second support d'enregistrement (1b).

6. Appareil d'enregistrement selon la revendication 5, adapté pour enregistrer ledit signal numérique sur ledit second support d'enregistrement (1b), comportant une plaque de base présentant un diamètre non supérieur à 120 millimètres et une épaisseur supérieure à 0,6 millimètre.

7. Appareil d'enregistrement selon la revendication 6, pour enregistrer des pistes sur le second support d'enregistrement (1b) selon un pas de piste d'approximativement 0,74 micromètre, selon une capacité d'enregistrement d'information non inférieure à 4,7 gigaoctets.

8. Appareil de reproduction pour reproduire un premier signal numérique qui est formaté au moyen de mots de données à multiples bits ou multibits qui sont échantillonnés à une première fréquence d'échantillonnage (f_s) à partir d'un premier support d'enregistrement (1a) et un second signal numérique qui est formaté au moyen de mots de données d'un bit qui sont échantillonnés à une seconde fréquence d'échantillonnage ($64f_s$) supérieure à la première fréquence d'échantillonnage (f_s) à partir d'un second support d'enregistrement (1b), comprenant :

une tête de lecture (21) pour reproduire un signal numérique à partir de l'un des premier et second supports d'enregistrement (1a, 1b) ;

un premier moyen de décodage (28) pour décoder les mots de données multibits échantillonnés à la première fréquence d'échantillonnage (f_s) en correspondance avec le premier support d'enregistrement (1a) ;

un second moyen de décodage (29) pour décoder les mots de données d'un bit échantillonnés à la seconde fréquence d'échantillonnage ($64f_s$) en correspondance avec le second support d'enregistrement (1b) ;

un moyen de discrimination (22) pour discriminer le premier support d'enregistrement (1a) et le second support d'enregistrement (1b) ; et

un moyen de commutation (65) pour commuter des signaux de sortie numériques en provenance du premier moyen de décodage (28) et du second moyen de décodage (29) conformément à un résultat dudit moyen de discrimination (22).

9. Appareil de reproduction selon la revendication 8, comprenant en outre :

un moyen de sur-échantillonnage (63) pour convertir la fréquence d'échantillonnage du signal de sortie numérique en provenance du premier moyen de décodage (28) selon la seconde fréquence d'échantillonnage (64fs) ; et un moyen de modulation delta-sigma pour convertir des mots de données multibits qui sont échantillonnés à la première fréquence d'échantillonnage (fs) selon des mots de données d'un bit qui sont échantillonnés à la seconde fréquence d'échantillonnage (64fs).

10. Appareil de reproduction selon la revendication 8 ou 9, comprenant en outre :

un moyen de sous-échantillonnage pour convertir la fréquence d'échantillonnage du signal de sortie numérique en provenance du second moyen de décodage (29) selon la première fréquence d'échantillonnage (fs).

11. Appareil de reproduction selon la revendication 8, 9 ou 10, dans lequel ledit premier moyen de décodage (28) décode des signaux en provenance dudit premier support d'enregistrement (1a) qui sont formatés au moyen de mots de données de 16 bits échantillonnés à 44,1 kHz.

12. Appareil de reproduction selon la revendication 8, 9, 10 ou 11, dans lequel ledit second moyen de décodage (29) décode des signaux en provenance dudit second support d'enregistrement (1b) qui sont formatés au moyen de mots de données d'un bit échantillonnés à 2,8224 MHz.

13. Appareil de reproduction selon la revendication 12, adapté pour reproduire ledit second signal numérique à partir dudit second support d'enregistrement (1b), comportant une plaque de base présentant un diamètre non supérieur à 120 millimètres et une épaisseur supérieure à 0,6 millimètre.

14. Appareil de reproduction selon la revendication 13, pour reproduire des pistes à partir du second support d'enregistrement (1b) qui sont formatées selon un pas de piste d'approximativement 0,74 micromètre et moyennant une capacité d'enregistrement d'information non inférieure à 4,7 gigaoctets.

15. Appareil de reproduction selon l'une quelconque des revendications 8 à 14, pour reproduire un support d'enregistrement (1b) sur lequel est enregistré un train de données d'un bit haute vitesse qui est émis en sortie depuis un convertisseur analogique-numérique à modulation delta-sigma à une fréquence d'échantillonnage de n fois 44,1 kHz, où n est un entier.

16. Appareil de reproduction pour reproduire un premier signal numérique qui est constitué par des premiers mots de données à multiples bits ou multibits qui sont échantillonnés à une première fréquence d'échantillonnage (fs) à partir d'un premier support d'enregistrement (1a) et un second signal numérique qui est constitué par un second mot de données à multiples bits ou multibits qui est échantillonné à une seconde fréquence d'échantillonnage (64fs) supérieure à la première fréquence d'échantillonnage (fs) à partir d'un second support d'enregistrement (1b), comprenant :

un moyen de reproduction (21) pour reproduire le signal numérique à partir d'un support pris parmi le premier support d'enregistrement (1a) et le second support d'enregistrement (1b) ;
un moyen de discrimination (22) pour discriminer le premier support d'enregistrement (1a) et le second support d'enregistrement (1b) ;
un premier moyen de décodage (28) pour décoder les données numériques en provenance du premier support d'enregistrement (1a) ;
un second moyen de décodage (29) pour décoder les données numériques en provenance du second support d'enregistrement (1b) ;
un premier moyen de génération d'horloge (60) pour générer un premier signal d'horloge (CK1) sur la base d'un signal d'entraînement en rotation pour le premier support d'enregistrement (1a) ;
un second moyen de génération d'horloge (61) pour générer un second signal d'horloge (CK2) sur la base d'un signal d'entraînement en rotation pour le second support d'enregistrement (1b) ;
un moyen de commutation (62) pour commuter les premier et second signaux d'horloge (CK1, CK2) conformément à un résultat du moyen de discrimination (22) ; et
un moyen de rotation (26) pour faire tourner le support d'enregistrement (1a, 1b) conformément au signal d'horloge qui est émis en sortie depuis le moyen de commutation (62).

17. Appareil de reproduction selon la revendication 16, comprenant en outre :

un moyen de conversion de fréquence

d'échantillonnage (63) pour convertir un nombre de bits et une fréquence d'échantillonnage du signal numérique en provenance du premier moyen de décodage (28) ; et

un second moyen de commutation (65) pour commuter le signal de sortie en provenance du moyen de conversion de fréquence d'échantillonnage (63) et le signal numérique en provenance du second moyen de décodage (29).

18. Appareil de reproduction selon la revendication 16, comprenant en outre :

un second moyen de commutation (65) pour commuter le signal de sortie en provenance du premier moyen de décodage (28) et le signal de sortie en provenance du second moyen de décodage (29) afin de former un signal de lecture qui est émis en sortie depuis l'appareil.

10

15

20

25

30

35

40

45

50

55

FIG. 1

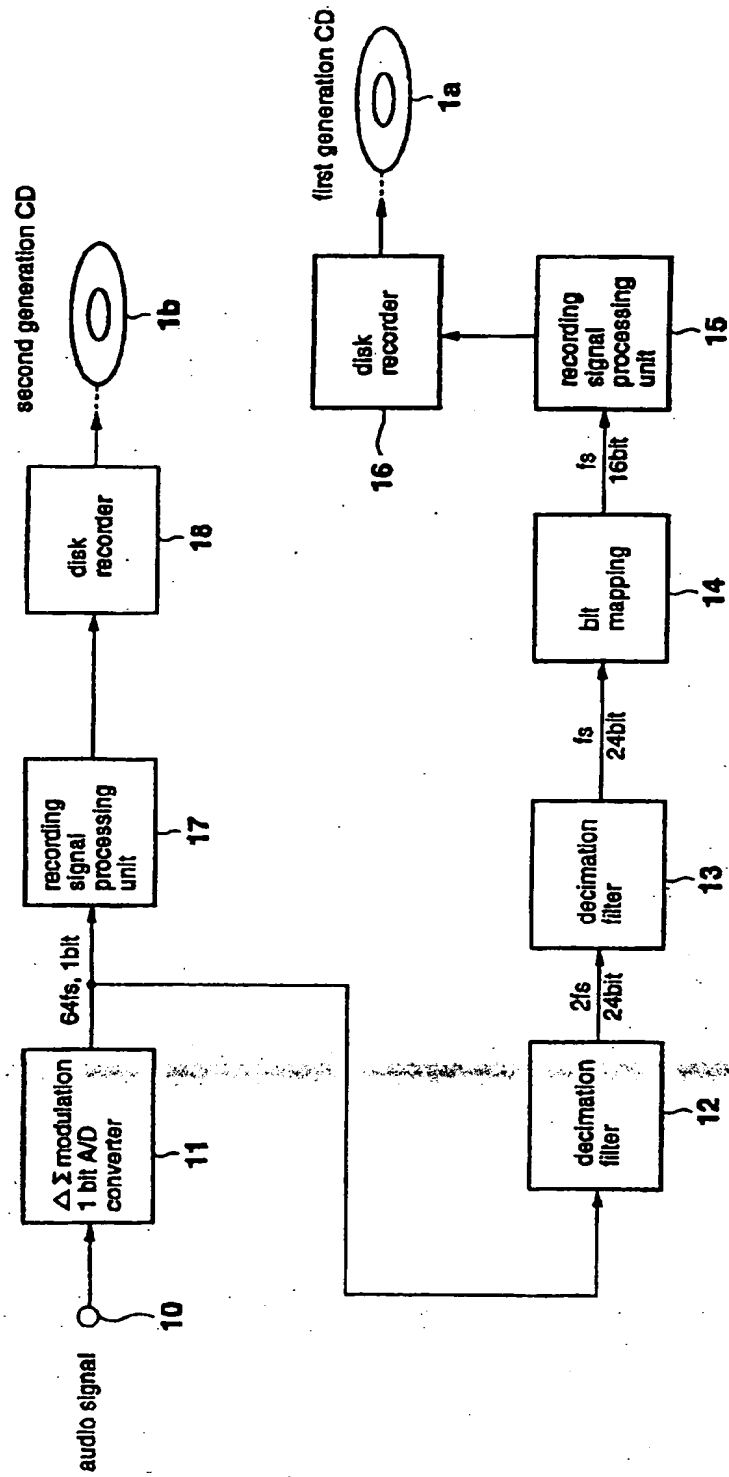


FIG. 2A

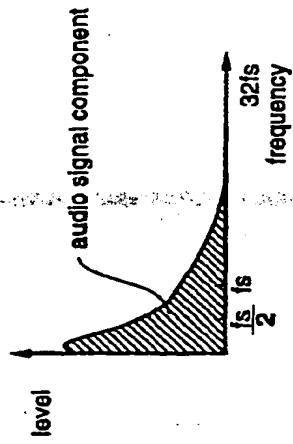


FIG. 2C

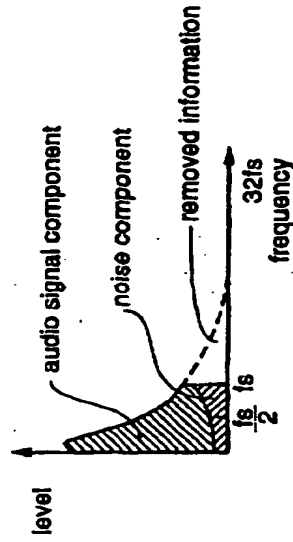


FIG. 2B

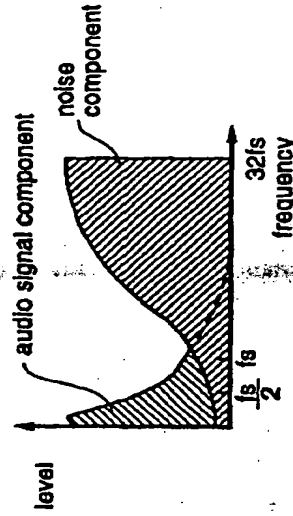


FIG. 2D

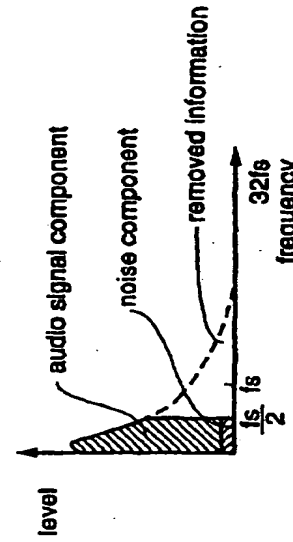


FIG. 3

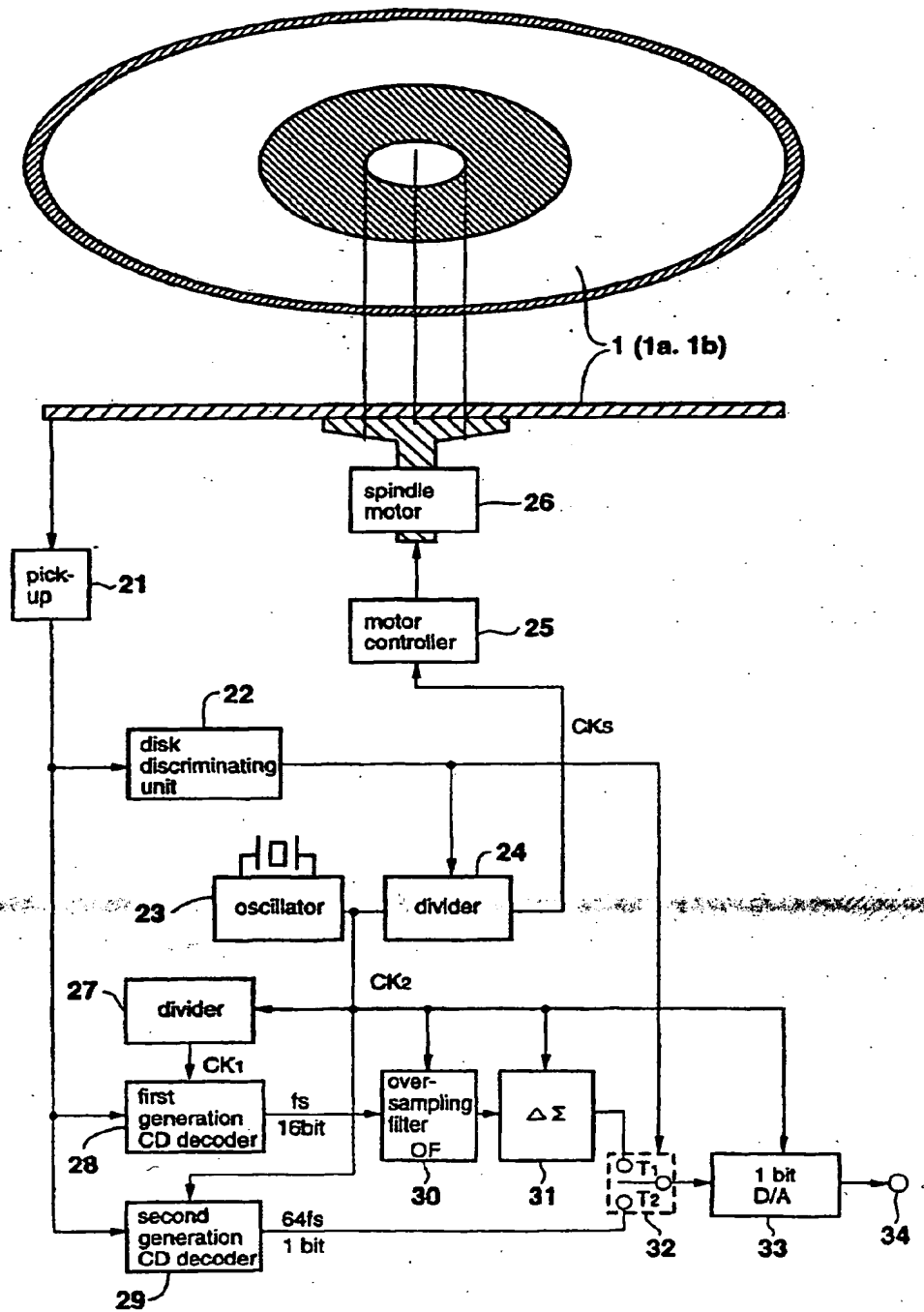


FIG. 4

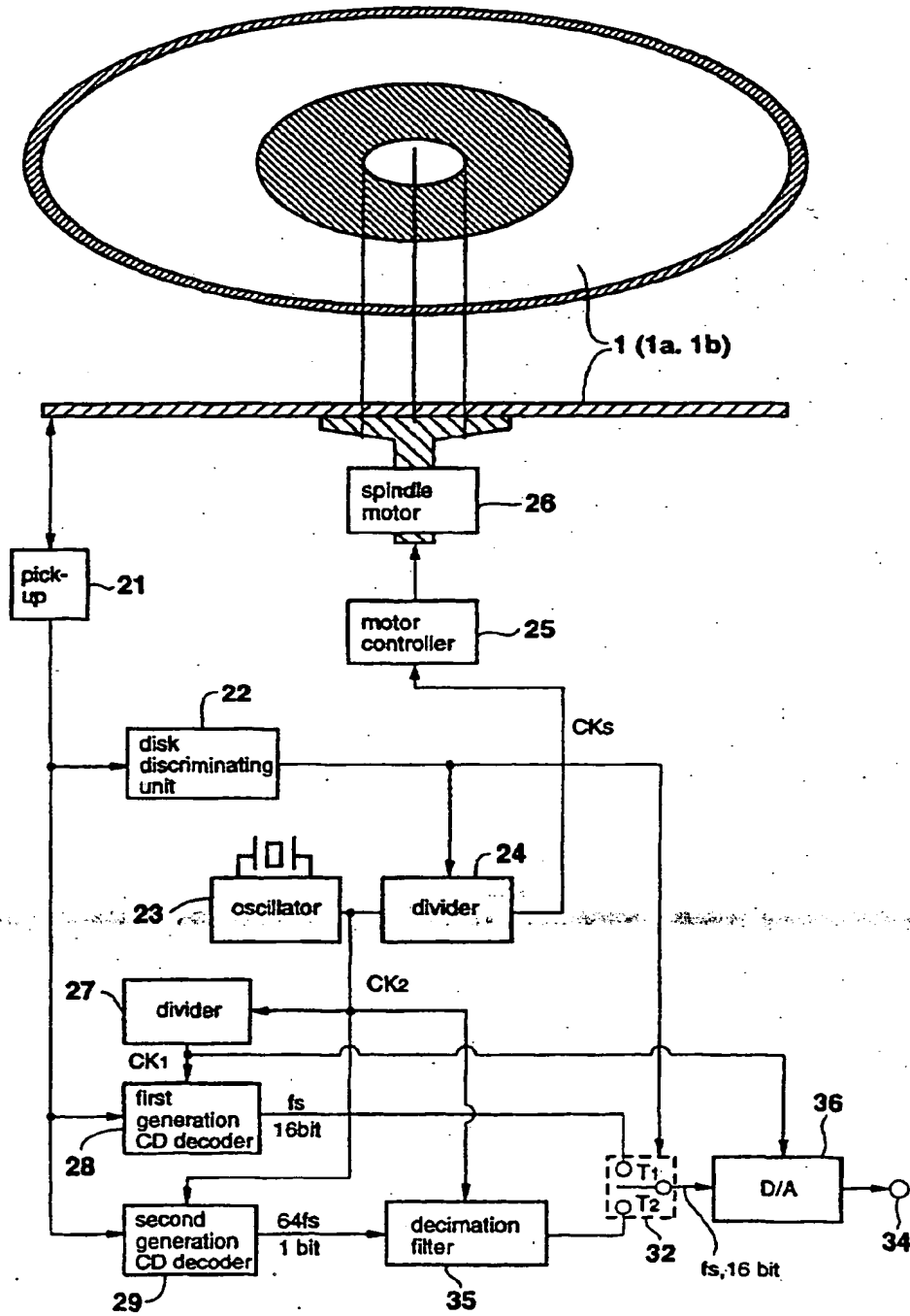


FIG. 5

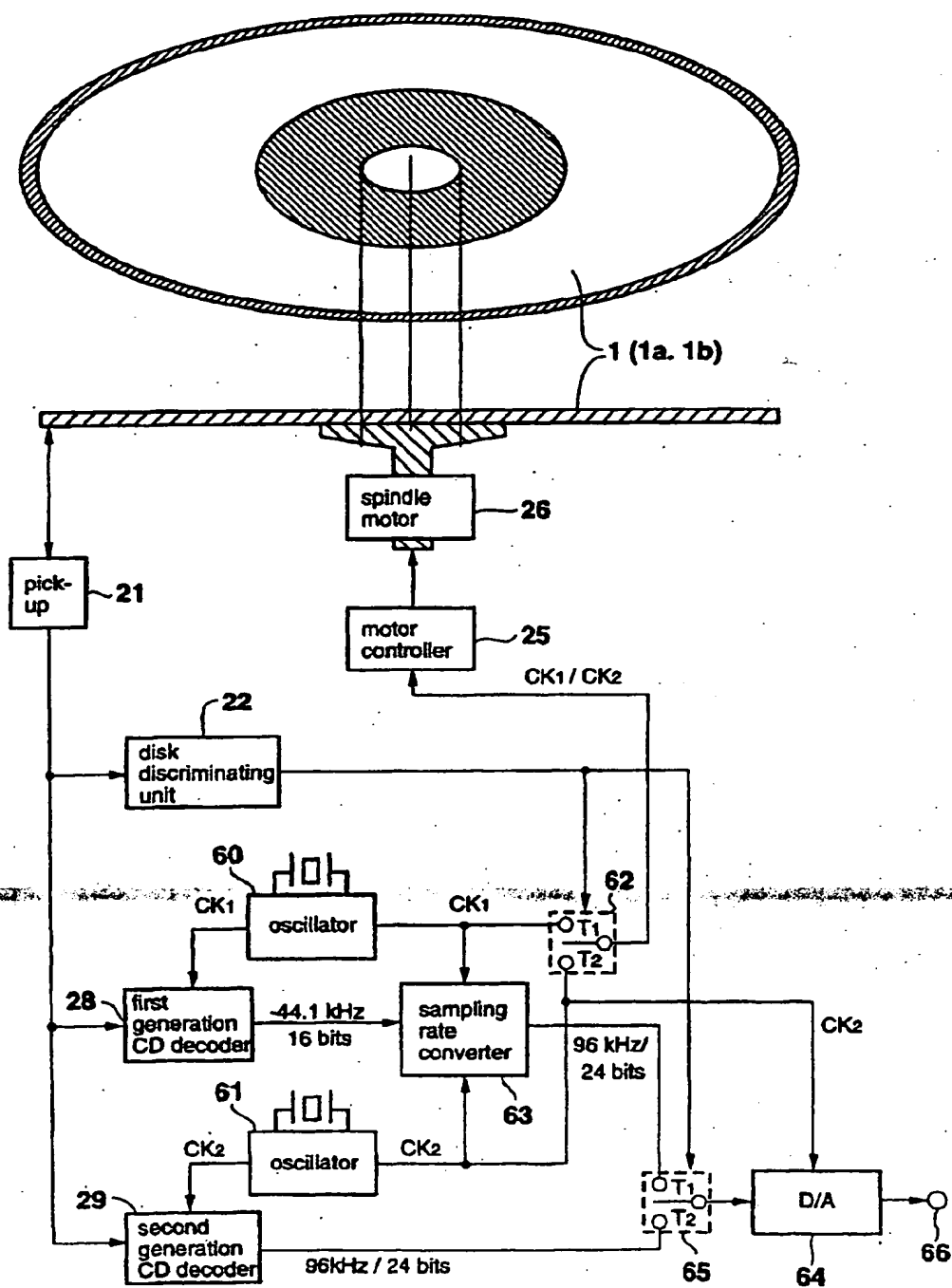


FIG. 6 (PRIOR ART)

